

WHAT IS CLAIMED IS:

- 1 1. A method for converting a signal to differing sample rates, comprising:
3 first signal;
4 operating on said plurality of data points to associate said signal with a
5 predetermined set of parameters, with said set of parameters including a first transition band
6 having an image corresponding thereto; and
7 varying said sample rate associated with said first signal by interpolation with
8 an interpolator having associated therewith a second transition band, with the width
9 associated with said second transition band being a function of a spectral separation of said
10 first transition band and said image, wherein a second signal is produced having a sequence
11 of data samples approximating the first signal.
- 1 2. The method recited in claim 1 wherein varying said sample rate
2 includes producing each data sample associated with said second signal by convolving a
3 predetermined finite number N of data points with an equal number of coefficients, with N
4 being greater than two.
- 1 3. The method recited in claim 2 wherein coefficients vary as a function
2 of the temporal spacing between the output point and the corresponding input points.
- 1 4. The method as recited in claim 1 wherein varying said sample rate
2 increases said sample rate.
- 1 5. The method as recited in claim 1 wherein varying said sample rate
2 decreases said sample rate.
- 1 6. The method as recited in claim 1 wherein operating on said plurality of
2 data points includes up-sampling said plurality of data points by a factor of two.
- 1 7. The method as recited in claim 1 wherein operating on said plurality of
2 data points includes filtering said plurality of data points with a half-band filter.
- 1 8. The method as recited in claim 1 wherein operating on said plurality of
2 data points includes decimating said plurality of data points with a half-band decimator.

1 9. The method as recited in claim 6 further including decimating a
2 plurality of data points output by said interpolator with a half-band decimator, with varying
3 said sample rate occurring after receiving said plurality of data points and before decimating
4 said plurality of data points.

1 10. The method as recited in claim 1 wherein operating on said plurality of
2 data points to associate said signal includes filtering the same with a finite impulse response
3 filter.

1 11. The method as recited in claim 1 wherein operating on said plurality of
2 data points to associate said signal includes filtering the same with an infinite impulse
3 response filter.

1 12. A method for converting a digital audio signal to a different sample
2 rate, comprising:
3 receiving a plurality of data points, associated with an audio signal, at an
4 initial sample rate;
5 halfband filtering said plurality of data points with a halfband filter; and
6 interpolating with an interpolator having independently programmable
7 parameters.

1 13. The method as recited in claim 12 wherein:
2 said halfband filtering is done in conjunction with upsampling said plurality of
3 data points; and
4 said interpolating follows said upsampling and halfband filtering.

1 14. The method as recited in claim 12 wherein:
2 said halfband filtering is done, without upsampling, on said plurality of
3 datapoints; and
4 said interpolating follows said halfband filtering.

1 15. The method as recited in claim 12 wherein:
2 said halfband filtering follows said interpolating.

1 16. The method as recited in claim 12 wherein:
2 said halfband filtering is done in conjunction with upsampling said plurality of

3 data points;

4 said interpolating follows said halfband filtering; and
5 halfband filtering and decimating following said interpolating.

1 17. A computer program product for converting signals to differing sample
2 rates comprising:

3 code for receiving a plurality of data points, associated with a signal, at a first
4 sample rate;

5 code for operating on said plurality of data points to associate said signal with
6 a predetermined set of parameters, with said set of parameters including a first transition band
7 having a first width;

8 code for varying said sample rate associated with said first signal by
9 interpolating a subset of data points of said plurality of data points with an interpolator
10 having associated therewith a second transition band, with the width associated with said
11 second transition band being a function of a spectral separation of said first transition band
12 and said image, wherein a second signal is produced having a sequence of data samples
13 approximating the first signal; and

14 a computer-readable storage medium for storing code.

1 18. The computer program product as recited in claim 17 wherein code for
2 operating on said plurality of data points includes code for up-sampling said plurality of data
3 points by a factor of two.

1 19. The computer program product as recited in claim 17 wherein code for
2 operating on said plurality of data points includes code for filtering said plurality of data
3 points with a half-band filter.

1 20. The computer program product as recited in claim 17 wherein code for
2 operating on said plurality of data points includes code for decimating said plurality of data
3 points with a half-band decimator.

1 21. The computer program product as recited in claim 18 further including
2 code for decimating said plurality of data points with a half-band decimator.

1 22. The computer program product as recited in claim 17 wherein code for
2 operating on said plurality of data points to associate said signal includes code for filtering

3 said data points with a filter selected from the set of filters consisting essentially of a finite
4 impulse response filter and a infinite impulse response filter.

1 23. A computer program product for converting a digital audio signal to a
2 different sample rate, comprising:
3 a computer-readable storage medium for storing code, said code including
4 code for receiving a plurality of data points, associated with an audio signal, at
5 an initial sample rate;
6 code for halfband filtering said plurality of data points with a halfband filter;
7 and
8 code for interpolating with an interpolator having independently
9 programmable parameters.

1 24. The computer program product as recited in claim 23 wherein:
2 said code for halfband filtering is executable in conjunction with code for
3 upsampling said plurality of data points; and
4 said code for interpolating is executable following said upsampling and
5 halfband filtering code.

1 25. The computer program product as recited in claim 23 wherein:
2 said code for halfband filtering is executable, without upsampling code, on
3 said plurality of datapoints; and
4 said code for interpolating is executable following said halfband filtering code.

1 26. The computer program product as recited in claim 23 wherein:
2 said code for halfband filtering is executable following said code for
3 interpolating.

1 27. The computer program product as recited in claim 23 wherein:
2 said code for halfband filtering is executable in conjunction with code for
3 upsampling said plurality of data points;
4 said code for interpolating is executable following said code for halfband
5 filtering; and further comprising:
6 code for halfband filtering and decimating executable following said code for
7 interpolating.

1 28. The method of claims 1 or 12 wherein said interpolator is an FIR Nth
2 order sum of products interpolator with linear interpolation of coefficients.

1 29. The computer program product of claims 17 or 23 wherein said
2 interpolator is an FIR Nth order sum of products interpolator with linear interpolation of
3 coefficients.

1 30. The method of claims 1 or 12 wherein said interpolator has a transition
2 band beginning adjacent the top of a passband and ending adjacent the bottom of a passband
3 image.

1 31. The computer program product of claims 17 or 23 wherein said
2 interpolator has a transition band beginning adjacent the top of a passband and ending
3 adjacent the bottom of a passband image.

1 32. The method of claims 7, 8, 9 or 12 wherein said halfband filter is an
2 IIR filter composed of first order allpass blocks.

1 33. The computer program product of claims 19 or 23 wherein said
2 halfband filter is an IIR filter composed of first order allpass blocks.
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